DEEP ENCODED LINGUISTIC AND ACOUSTIC CUES FOR ATTENTION BASED END TO END SPEECH EMOTION RECOGNITION



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Introduction

- Speech Emotion Recognition (SER) has several applications
 - man-machine interactions
 - human health assistance
 - call center analytics etc.

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 - cross-domain knowledge transfer have significantly impacted SER.

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 - man-machine interactions
 - human health assistance
 - call center analytics etc.
- Developments in deep learning especially in terms of,
 - data augmentation
 - better feature extractors
 - cross-domain knowledge transfer have significantly impacted SER.
- Can be further improved by exploiting,
 - Acoustic information : Spectrograms from raw audio and glottal source signals
 - Linguistic information : Text, Phoneme sequences, intermediate DNN representations

Two directions :

- Use complex hand-crafted features (ex: OpenSMILE feature set)
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Transferring knowledge within tasks/datasets^[1]

- In Deep networks,
 - □ initial layers \rightarrow low-level features
 - □ final layers \rightarrow high-level features
- Transfer learning → share knowledge across datasets and tasks.



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"Low-level features are more generic and easier to transfer from one task to another" Could there be exceptions?

Jointly learning supplementary tasks ^[2]

- Uncertainty about most relevant and robust features/layers
- Progressive network : training ASR and SER tasks jointly
- ASR representations show improved performance mainly due to the robustness to speaker and condition variations.



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Key Takeaways from related work:

- Influence of linguistic knowledge in spoken utterances for SER task still remains unexplored.
- Selection of intermediate ASR layers needs to be studied thoroughly.

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Proposed System

Representative features

Acoustic features : Mel-spectrogram

- Sampling rate = 16 kHz
- Frame duration = 25 msec
- Length of FFT window = 2048
- Hop length = 400 samples
- Number of bins on mel-scale = 128

Concatenate Δ and Δ - Δ for the mel-spectrogram.

Representative features

Deep encoded Linguistic features : DeepSpeech ASR [3]

Note : Layers closer to output capture the linguistic content of speech while the layers close to input capture the acoustic content.^[4]



DeepSpeech–1 architecture

[3] Mozilla, "DeepSpeech–0.4.0,"https://github.com/mozilla/DeepSpeech/releases, January 2019
 [4] Swapnil B, Imran S, Sunil K, "End-to-End spoken language understanding: Bootstrapping in low resource scenarios," Interspeech 2019.

Deep encoded Linguistic and Acoustic cues for SER

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Can we get linguistic context of embedded emotion in the spoken utterance?



DeepSpeech-1 architecture

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Representative features



Visualization of activations from different layers of DeepSpeech model, for the same utterance spoken in different emotions. Columns represent the 6 layers and rows represent emotions. *anger; fearful; happy; calm; sac*

- 1st, 2nd and 5th layers show least correlation across the rows (emotions).
- Lesser correlation in 1st and 2nd layer is due to variations in speaker, gender etc.^[4]
- We use the output from the 5th layer for getting the linguistic context for the SER task.

Proposed architecture

Encoder :

- 2 layers of 1-D convolutions.
 - Helps to learn temporal context between adjacent frames.
- 1-D convolution layer
 Retain normalization
 - → Batch normalization layer
 - → ReLU activation

Decoder :

- Multi-head self attention layer
 - ➔ Average pooling
 - ➔ 2 feedforward dense layers.

Output :

Softmax distribution over individual emotions.



Proposed Encoder – Decoder model architecture

Multi-head Self Attention

- Let *E* be the output of the encoder block
- *W_i* are trainable weight matrices
- d_i is the dimension
- A_i : Attention weight of a single head
- A_{MH} : Final multi-head self attention
- *h* : total number of heads

$$A_{i} = softmax \left(\frac{Q_{i}K_{i}^{T}}{\sqrt{d_{K}}}\right) V_{i} \forall i \in \{1, 2, \dots h\}$$
$$Q = EW_{Q}, \quad K = EW_{K}, \quad V = EW_{V}$$
$$A_{MH} = (A_{1} ||A_{2}|| \dots ||A_{h}) W_{E}$$
$$Context, \ C = E + A_{MH}$$



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Var.

 E_i

 Q_i

 K_i

 $Q_i K_i^T$

 V_i

 $(Q_i K_i^T) V_i$

Dim.

Experiments

- Dataset : IEMOCAP^[5]
 - Recording setups :
 - Categorical Emotion classes :
- Model configurations :

- 2 { Improvised speech, scripted play }
 4 { opport borning on a suttral opdages
- 4 { anger, happiness, neutral, sadness }



[5] Carlos Busso, IEMOCAP: Interactive Emotional Dyadic Motion Capture database," Language resources and evaluation, 2008.

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Deep encoded Linguistic and Acoustic cues for SER

Experiments with improvised recordings

Model (Input features)	Weighted Acc., WA	Unweighted Acc., UA
Yenigalla et al.,2018 [6] (only spectrogram)	71.3	61.6
Satt et al., 2017 [7]	68.8	59.4
Lee et al., 2015 [8]	63.8	62.85
Model - 1 (acoustic)	72.08	58.53
Model - 1 (downsampling + ensembling)	70.05	63.27
Model - 2 (linguistic)	69.56	54.62
Model - 3 (fusion)	72.34	58.31

Observation :

- Improvement using only Acoustic features
- Improvement using Linguistic features (or +Acoustic features) ×

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<u>Reasoning</u> : Improvised recordings carry less linguistic correlations and capture emotion representative characteristics mostly in acoustic space.

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What if there is <u>linguistic context embedded</u> within the samples?

Experiments with scripted recordings

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> 7.64% improvement compared to "only acoustic features" as input.

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What if the data itself has a combination of both scripted and improvised speech ?

Experiments with scripted + improvised recordings

Model (Input features)	Weighted Accuracy, WA	Unweighted Accuracy, UA
Model - 1 (acoustic)	70.82	55.53
Model - 2 (linguistic)	62.03	51.96
Model - 3 (fusion)	65.05	58.39
Model - 3 (downsampling + ensembling)	68.11	63.15

Observation :

- Improvement using Acoustic + Linguistic features ✓
- Improvement using Acoustic features or only Linguistic features

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- Class imbalance in the combined scenario plays important role
 - Model -1 achieves best WA but very low UA
- Fusion of linguistic information + acoustic features -> + 2.89% in UA

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But, is the self-attention module actually helping?

Discussion

$$A_i = softmax \left(\begin{array}{c} Q_i K_i^T \\ \sqrt{d_K} \end{array} \right) V_i$$

- Model learns the acoustically significant frames and weighs them heavily during the formation of context.
- Strong emphasis around the word "everything" makes it almost distinctive as anger emotion.
- Not all heads contribute equally, most important and confident heads play a consistent role.



Attention weights (a T X T matrix) for each attention head. T : timesteps, True emotion : *anger*

Conclusion

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- Less correlation of linguistic cues with the emotion than its acoustic counterpart in the improvised recordings.
- Combination of linguistic and acoustic features gives an improvement of
 - 6.29% for only scripted
 - 2.86% for combined scenario indicating usefulness of our approach.

Thank You !

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